

DEFENSE INFORMATION SYSTEMS AGENCY

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 $\frac{\text{IN REPLY}}{\text{REFER TO:}}$ Joint Interoperability Test Command (JTE)

10 Aug 12

MEMORANDUM FOR DISTRIBUTION

SUBJECT: Extension of the Special Interoperability Test Certification of Interactive

Intelligence[®] Inc. from Customer Interaction CenterTM (CIC) Release 3.0 Software

Update (SU) 14 to CIC Release 4.0 SU2

References: (a) DoD Directive 4630.05, "Interoperability and Supportability of Information Technology (IT) and National Security Systems (NSS)," 5 May 2004

(b) CJCSI 6212.01E, "Interoperability and Supportability of Information Technology and National Security Systems," 15 December 2008

(c) through (g), see Enclosure

- 1. References (a) and (b) establish the Defense Information Systems Agency (DISA), Joint Interoperability Test Command (JITC), as the responsible organization for interoperability test certification.
- 2. The Interactive Intelligence® Inc. CIC Release 3.0 SU9 is hereinafter referred to as the system under test (SUT). The SUT meets all of its critical interoperability requirements and is certified for joint use within the Defense Information System Network (DISN) as a Private Branch Exchange (PBX) 2. The PBX2 switches have no Military Unique Features (MUFs) and can only serve Department of Defense (DoD), non-DoD, non-governmental, and foreign government users having no missions or communications requirement to ever originate or receive Command and Control (C2) communications. Since PBX2s do not support MUF requirements detailed in Reference (c), connectivity to the DSN is not authorized until a waiver is granted by the Chairman of the Joint Chiefs of Staff (CJCS) for each site in accordance with Reference (d). No other configurations, features, or functions, except those cited within this report, are certified by the JITC. This certification expires upon changes that could affect interoperability, but no later than 5 April 2014, which is three years from the date the DISA Field Security Operations (FSO) provided a positive Certification and Accreditation (CA) recommendation for the original certification.
- 3. The extension of this certification is based upon Desktop Reviews (DTRs) 4 and 9. The original certification is based on interoperability testing, review of the vendor's Letters of Compliance (LoC), and FSO CA recommendation. Interoperability testing of the SUT was conducted at JITC's Global Information Grid Network Test Facility at Fort Huachuca, Arizona, from 22 through 30 July 2010 and documented in Reference (e). Review of vendor's LoC was completed on 30 March 2011. The FSO provided a positive CA recommendation on 5 April 2011 based on the security testing completed by DISA-led IA test teams and published in a separate report, Reference (f). The requested DTR updates are discussed in Table 1. JITC

determined these DTRs required verification and validation (V&V) testing. JITC conducted V&V testing for DTRs 4 and 9 from 5 through 23 December 2011 and 30 April through 11 May 2012. The DISA CA approved the IA posture for DTRs 4 and 9 on 11 July 2012. Based on the results of this testing and the DISA CA approval of the IA posture for these DTRs, JITC approves these DTRs.

Table 1. SUT DTR Descriptions

I	Interactive Intelligence [®] Inc. Customer Interaction Center TM (CIC) Release 4.0 Software Updates							
DT	ΓR	DTR Description						
4	4	This DTR updates the SUT from software version CIC 3.0 SU14 to CIC 4.0 SU1. SU1 includes periodic maintenance						
		updates. Additionally, this DTR includes the	following:					
		 New Audio Codes M800 Line Side and T 	Trunk Side Gatew	ays.				
		 Upgrades the server to Windows 2008 R2 	2 64 bit and client	ts to Windows 7 64 bit.				
		Adds Polycom Sound Point IP IPv4 only	end instruments.					
		 Upgrades system to IPv6 dual stack capal 	ble.1					
9	9	This DTR updates the SUT from Software Vo	ersion CIC 4.0 SU	U1 to CIC 4.0 SU2, which includes periodic maintenance				
		updates.						
NOTE:								
1. The 5	SUT is IP	v6 dual stack capable; however, the IP end inst	truments are not.	DISA adjudicated this as minor on 22 May 2012 with the				
vendor's	s POA&N	If to support IPv6 EIs in new software update to	argeted for release	e in December 2012. The SUT will be subject to removal				
from the	e APL if t	he POA&M is not met.						
LEGEN								
CIC	Custo	tomer Interaction Center R2 Release 2		Release 2				
DTR	Deskt	op Review	SU	Software Update				
IP		et Protocol	System Under Test					
IPv6	Intern	et Protocol version 6						
l								

- 4. The interoperability test summary of the SUT is indicated in Table 2. The PBX2 Capability Requirements (CRs) and Feature Requirements (FRs) are listed in Table 3. This interoperability test status is based on the SUT's ability to meet:
 - a. DSN services for Network and Applications specified in Reference (c).
- b. PBX2 interface and signaling requirements for trunks/lines specified in Reference (d) verified through JITC testing and/or vendor submission of LoC.
- c. PBX2 CRs/FRs specified in Reference (d) verified through JITC testing and/or vendor submission of LoC.
- d. The overall system interoperability performance derived from test procedures listed in Reference (g).

Table 2. SUT Interoperability Test Summary

DSN Trunk Interfaces						
Interfac	e & Signaling	Critical	Status	Remarks		
T1 C	AS (DTMF)	No ¹	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
E1 CAS (DTMF)		No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
(AN	ON PRI NI 1/2 ISI T1.607)	No¹	Certified	Met all critical CRs and FRs however is not required for a PBX2. ²		
	ISDN PRI J-T Q.931)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
DSN Line Interfaces						
	e & Signaling	Critical	Status	Remarks		
(GR-	ire Analog 506-CORE)	Yes	Certified	Met all critical CRs and FRs.		
	BRI NI 1/2	No	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
	VoIP et IEEE 802.3u)	No No	Not Tested Certified	This interface is not supported by the SUT and is not required for a PBX2. Met all critical CRs and FRs with the following minor exception: The Polycom Sound Point IP EI are IPv4 only and are not certified for shared access. ³		
			DSN Featu	res and Capabilities		
Featur	e/Capability	Critical	Status	Remarks		
Common Features		Yes	Certified	Met all critical CRs and FRs.		
A	Attendant		Certified	Met all critical CRs and FRs.		
Pul	Public Safety		Certified	Met all critical CRs and FRs for the basic 911.		
Call	Call Processing		Certified	Met all critical CRs and FRs.		
ISD	N Services	No	Certified	Met all critical CRs and FRs with the T1 ISDN PRI interface.		
Sync	chronization	Yes	Certified	Met all critical CRs and FRs.		
Vo	IP System	No	Certified	Met all critical CRs and FRs. ⁴		
	Security	Yes	Certified	See note 5.		
			Netw	ork Gateways		
Gateway	Interface & Signaling	Critical	Status	Remarks		
	T1 CAS (DTMF)	No ¹	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
	E1 CAS (DTMF)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
PSTN	T1 ISDN PRI NI 1/2 (ANSI T1.607)	No ¹	Certified	Met all critical CRs and FRs.		
	E1 ISDN PRI (ITU-T Q.931)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
	Ground Start Line (GR-506-CORE)	No	Not Tested	This interface is not supported by the SUT and is not required for a PBX2.		
	Loop Start Line (GR-506-CORE)	No	Certified	Met all critical CRs and FRs with an LoC.		

Table 2. SUT Interoperability Test Summary (continued)

NOTES:

- 1. The UCR 2008 does not specify a required interface for a PBX2. A PBX2 switch must support at least T1 CAS or T1 PRI.
- 2. The vendor included the Audiocodes M800 T1 PRI with Patch 6.20A.059 as part of the SUT to accommodate secure call capability.
- 3. The SUT Polycom Sound Point IP EIs only support IPv4. This was adjudicated by DISA on 22 May 2012 as minor with the vendors Plan of Action and Milestone to support Polycom IPv6 EIs in new software targeted for release in December 2012.. The Polycom IP EIs have the capability of shared access (data connection); however, this was not tested and is not certified for joint use. Shared access is conditional for a PBX2.
- 4. The SUT is IPv6 dual stack capable; however, the IP end instruments are not. DISA adjudicated this as minor on 22 May 2012 with the vendor's POA&M to support IPv6 EIs in new software update targeted for release in December 2012. The SUT will be subject to removal from the APL if the POA&M is not met.
- 5. Security is tested by DISA-led Information Assurance test teams and published in a separate report, Reference (f).

Standard for carrier sense multiple access with	ITU-T	International Telecommunication Union -
collision detection at 100 Mbps		Telecommunication Standardization Sector
American National Standards Institute	LoC	Letters of Compliance
Assistant Secretary of Defense for Networks and	LSSGR	Local Access and Transport Area (LATA) Switching
Information Integration		Systems Generic Requirements
Basic Rate Interface	Mbps	Megabits per second
Channel Associated Signaling	NI 1/2	National ISDN Standard 1 or 2
Capability Requirements	OAM	Operations Administration and Management
Defense Information Systems Agency	PBX2	Private Branch Exchange 2
Defense Switched Network	PRI	Primary Rate Interface
Digital Subscriber Signaling 1	PSTN	Public Switched Telephone Network
Dual Tone Multi-Frequency	Q.931	Signaling Standard for ISDN
European Basic Multiplex Rate (2.048 Mbps)	SUT	System Under Test
Feature Requirements	T1	Digital Transmission Link Level 1 (1.544 Mbps)
Generic Requirement	T1.607	ISDN Layer 3 Signaling Specification for Circuit Switched
LSSGR: Signaling for Analog Interfaces		Bearer Service for DSS1
Institute for Electrical and Electronics Engineers	UCR	Unified Capabilities Requirements
Internet Protocol version 6	VoIP	Voice over Internet Protocol
Integrated Services Digital Network		
	collision detection at 100 Mbps American National Standards Institute Assistant Secretary of Defense for Networks and Information Integration Basic Rate Interface Channel Associated Signaling Capability Requirements Defense Information Systems Agency Defense Switched Network Digital Subscriber Signaling 1 Dual Tone Multi-Frequency European Basic Multiplex Rate (2.048 Mbps) Feature Requirements Generic Requirement LSSGR: Signaling for Analog Interfaces Institute for Electrical and Electronics Engineers Internet Protocol version 6	collision detection at 100 Mbps American National Standards Institute Assistant Secretary of Defense for Networks and Information Integration Basic Rate Interface Channel Associated Signaling Channel Associated Signaling Channel Associated Signaling NI 1/2 Capability Requirements OAM Defense Information Systems Agency Defense Switched Network Digital Subscriber Signaling 1 PSTN Dual Tone Multi-Frequency European Basic Multiplex Rate (2.048 Mbps) SUT Feature Requirement Generic Requirement T1 Generic Requirement LSSGR: Signaling for Analog Interfaces Institute for Electrical and Electronics Engineers Internet Protocol version 6 VoIP

Table 3. PBX2 Requirements

DSN Trunk Interfaces					
Interface	Critical		Requirements Required or Conditional	References	
T1 CAS (MFR1, DTMF, DP)	No No (Europe only)	Trunking	Direct Inward Dialing (C) National ISDN 1/2 Primary Access (C) ITU-T ISDN Primary Access (Europe only) (C) Normal Wink Start Operations (C) Glare Operation (C) Abnormal Wink Start (C) Glare Resolution (C) Call for Service Timing (C) Guard Timing (C) Satellite Interface (C) Disconnect Control (C) Reselect and Retrial (C) Off-Hook Supervision Transition (C) Dial-Pulse Signals (C) DTMF Signaling (C) DSN ISDN User-to-Network Signaling (C) Application (C) Physical Layer (C) Data Link Layer (C) Data Link Connection (C) Peer-to-Peer Procedures of Data-Link Layer (C)	 UCR Section 5.2.1.3.2 UCR Section 5.2.1.3.4.1 UCR Section 5.2.1.3.4.2 UCR Section 5.2.4.3.3.1.1 UCR Section 5.2.4.3.3.1.2 UCR Section 5.2.4.3.3.2 UCR Section 5.2.4.3.3.2.2 UCR Section 5.2.4.3.5 UCR Section 5.2.4.3.6 UCR Section 5.2.4.3.7 UCR Section 5.2.4.3.8 UCR Section 5.2.4.3.9 UCR Section 5.2.4.3.10 UCR Section 5.2.4.4.1 UCR Section 5.2.4.4.1 UCR Section 5.2.4.7.1 UCR Section 5.2.4.7.1.1 UCR Section 5.2.4.7.1.2 UCR Section 5.2.4.7.1.3 UCR Section 5.2.4.7.1.3.1 UCR Section 5.2.4.7.1.3.1 UCR Section 5.2.4.7.1.3.2 	
T1 ISDN PRI NI 1/2 (ANSI T1.607)	No		 Layer 3 DSN User-to-Network Signaling (C) DSN User-to-Network Signaling for Circuit-Switched Bearer Services (C) Sequence of Messages for DSN Circuit-Switched Calls (C) Message Functional Definition and Content (C) General Message Format and Information Elements Coding (C) Supplementary Services (C) Transmission (R) PCM-24 Digital Trunk Interface (R) Interface Characteristics (R) Supervisory Channel Associated Signaling (C) Clear Channel Capability (C) Alarm and Restoral Requirements (C) PCM-30 Digital Trunk Interface (Europe only) (C) Interoperation of PCM-24 and PCM-30 (C) Analog Trunk Interface (C) 	 UCR Section 5.2.4.7.1.4.2 UCR Section 5.2.4.7.1.4.2 UCR Section 5.2.4.7.1.4.3 UCR Section 5.2.4.7.1.4.4 UCR Section 5.2.4.7.1.4.5 UCR Section 5.2.4.7.1.4.6 UCR Section 5.2.5 UCR Section 5.2.6.1 UCR Section 5.2.6.1.1 UCR Section 5.2.6.1.2 UCR Section 5.2.6.1.3 UCR Section 5.2.6.1.4 UCR Section 5.2.6.2 UCR Section 5.2.6.3 UCR Section 5.2.6.3 UCR Section 5.2.6.4 	
E1 ISDN PRI (ITU-T Q.931)	No (Europe only)	Voice	Integrated Digital Loop Carrier (C) MOS (R) Secure calls (C)	UCR Section 5.2.6.5 UCR 5.2.12.8.2.1 CJCSI 6215.01C	
		Pacsimile Data	 Analog: ITU-T T.4 (R) Modem (VBD) (R) 56 kbps switched data (C: PRI only) 64 kbps switched data (C: PRI only) NX56 synchronous BER (C: PRI only) NX64 synchronous BER (C: PRI only) Secure data (STE/STU-III) (C) 	DISR CJCSI 6215.01C UCR Section 5.2.2.9.6 CJCSI 6215.01C	
		VTC	• ITU-T H.320 (R: PRI only)	• FTR 1080B-2002	

Table 3. PBX2 Requirements (continued)

		DSN Line Interfaces	
Interface	Critical	Requirements	References
interrace	Critical	Required or Conditional	References
		Individual Line (R)	• UCR Section 5.2.1.1.1
		• PBX Line (C)	• UCR Section 5.2.1.3.1
2-Wire Analog	Yes	 National ISDN 1/2 Basic Access (C) 	• UCR Section 5.2.1.3.3
		Access • Analog Line (C)	• UCR Section 5.2.1.3.5
		• Loop Start Line (R: 2-Wire Analog only)	• UCR Section 5.2.4.2.1
ISDN BRI NI 1/2	No	Reverse Battery (C)	• UCR Section 5.2.4.3.1
		• S/T Reference Point (ISDN BRI) (C)	• UCR Section 5.2.4.7.1.2.1
0 W. D	N	Voice • MOS (R)	• CJCSI 6215.01C
2-Wire Proprietary	No	Secure Calls (C)	• CJCSI 6215.01C
Digital		Facsimile • Analog: ITU-T T.4 (R)	• DISR
		Modem (VRD) (R)	• CJCSI 6215.01C
		Data Secure data (STE/STU-III) (C)	• CJCSI 6215.01C
		VTC • ITU-T H.320 (R: BRI only)	• FTR 1080B-2002
	1	DSN Features & Capabilities	
Feature/		Requirements	
Capability	Critical	Required or Conditional	References
Cupubility		Individual Lines (R)	• UCR Section 5.2.4.7.1.2.1
		Call waiting (C)	• UCR Section 5.2.1.1.5.1
	Yes	Three-way calling (C) Three-way calling (C)	• UCR Section 5.2.1.1.6
		Add-on transfer, conference calling, and call hold (C)	 UCR Section 5.2.1.1.7
		Call Transfer Individual – All calls (C)	 UCR Section 5.2.1.1.7 UCR Section 5.2.1.1.7.1
		Call Transfer Individual – All Calls (C) Call Transfer - Internal Only (C)	• UCR Section 5.2.1.1.7.1
		Call Transfer - Individual – Incoming Only/Add-On	• UCR Section 5.2.1.1.7.3
		Consultation Hold – Incoming Call (C)	• UCR Section 5.2.1.1.7.3
		Call Transfer – Outside (C)	 UCR Section 5.2.1.1.7.3 UCR Section 5.2.1.1.7.4
Common Features		Call Transfer – Outside (C) Call Transfer – Add-On Restricted Station (C)	• UCR Section 5.2.1.1.7.5
		Call Transfer – Attendant (C) Call Transfer – Attendant (C)	• UCR Section 5.2.1.1.7.6
		• Call Hold (C)	 UCR Section 5.2.1.1.7.7
		Conference Calling – Six Way Station Controlled (C)	• UCR Section 5.2.1.1.7.8
		Call forwarding Variable (C)	• UCR Section 5.2.1.1.8.1
		Call Forward Busy Line (C)	• UCR Section 5.2.1.1.8.2
		Call Forwarding – Don't Answer – All Calls (C)	• UCR Section 5.2.1.1.8.3
		Selective Call Forwarding (C)	• UCR Section 5.2.1.1.8.4
		Call pick-up (C)	 UCR Section 5.2.1.1.8.4 UCR Section 5.2.1.1.9.1
Attendant	No	Attendant Features (C)	• UCR Section 5.2.1.2
Attendant	NO	\ \ \ \ \ \ \ \ \ \ \ \ \ \ \ \ \ \ \	
D 11' C C .	37	• Emergency Service (911) Caller (R)	• UCR Section 5.2.1.4.1.1
Public Safety	Yes	• Emergency Service (911) Public Safety Answering Service (C)	• UCR Section 5.2.1.4.1.2
		Enhanced Emergency Service (E911) (C)	• UCR Section 5.2.1.4.1.3
		• Origination Treatment (R)	• UCR Section 5.2.3.1.1
		Originating Busy (R)	• UCR Section 5.2.3.1.1.1
		• Termination Treatment (R)	• UCR Section 5.2.3.1.2
		Busy or Idle Status (C)	• UCR Section 5.2.3.1.2.1
		• Release Treatment (R)	• UCR Section 5.2.3.1.3
		• Interruption Treatment (R)	• UCR Section 5.2.3.1.4
C II D	37	• Connections (R)	• UCR Section 5.2.3.1.5
Call Processing	Yes	• Class of Service (C)	• UCR Section 5.2.3.1.6
		• E&M Lead Signaling States (C)	• UCR Section 5.2.3.3.1
		• 4-Wire Analog User Access Lines (C)	• UCR Section 5.2.3.3.2
		• 2-Wire User Access Lines (C)	• UCR Section 5.2.3.3.3
		• Interswitch and Intraswitch Dialing (C)	• UCR Section 5.2.3.5.1.2
		Calling Name Delivery (C)	• UCR Section 5.2.3.5.1.8.1
		Calling Number Delivery (C)	• UCR Section 5.2.3.5.1.8.2
		• Screening (C)	• UCR Section 5.2.3.5.8

Table 3. PBX2 Requirements (continued)

DSN Features & Capabilities (continued)					
Feature/ Capability	Critical	Requirements Required or Conditional	References		
ISDN Services	No	 BRI Access, Call Control and Signaling (C) Uniform Interface Configuration for BRIs (C) BRI Features (C) PRI Access, Call Control and Signaling (C) PRI Features (C) Packet Data Features and Capabilities (C) 	 UCR Section 5.2.9.2 Table 5.2.9.1 UCR Section 5.2.9.2 Table 5.2.9-2 UCR Section 5.2.9.2 Table 5.2.9-3 UCR Section 5.2.9.2 Table 5.2.9-4 UCR Section 5.2.9.2 Table 5.2.9-5 UCR Section 5.2.9.2 Table 5.2.9-6 		
Synchronization	Yes	 Line timing mode (C) Internal Stratum 4 (R) Synchronization Performance Monitoring Criteria (C) DS1 Traffic Interfaces (C) DS0 Traffic Interconnects (C) 	 UCR Section 5.2.10.1.1.2 UCR Section 5.2.10.1.2.2 UCR Section 5.2.10.2 UCR Section 5.2.10.3 UCR Section 5.2.10.4 		
Security	Yes	• GR-815, STIGs, and DoDI 8510.bb (DIACAP) (R)	• UCR Section 3		
VoIP System	No	VoIP function is conditional. If VoIP is provided, all of the following requirements must be met: • Voice Quality with MOS of 4.0 or better (R) • ITU-T G.711 PCM CODEC (R) • MLPP (C) • Security (R) • Network management (C) • System timing (R) • Latency ≤ 60 milliseconds (R) • IPv6 capable (R) • Service Class Tagging (R)	 UCR section 5.2.12.8.2.1 UCR section 5.2.12.8.2.2 UCR section 5.2.12.8.2.3 UCR section 5.2.12.8.2.4 UCR section 5.2.12.8.2.5 UCR section 5.2.12.8.2.6 UCR section 5.2.12.8.2.7 UCR section 5.2.12.8.2.8 UCR section 5.2.12.8.2.8 UCR section 5.2.12.8.2.9 		
		Network Gateways			
Gateway	Critical	Requirements Required or Conditional	References		
PSTN	No	Positive Identification Control (C) On-Netting (C) Off-Netting (C) Loop Start Line (C) Ground Start Line (C) Immediate Start (C) Delay Dial (C)	 CJCSI 6215.01C CJCSI 6215.01C CJCSI 6215.01C UCR Section 5.2.4.2.1 UCR Section 5.2.4.2.2 UCR Section 5.2.4.3.2 UCR Section 5.2.4.3.4 		

Table 3. PBX2 Requirements (continued)

LEGEND) :				
ANSI	American National	FTR	Federal Telecommunications	PCM-24	Pulse Code Modulation - 24
	Standards Institute		Recommendation		Channels
BER	Bit Error Ratio		Video Teleconferencing Services	PCM-30	Pulse Code Modulation - 30
BRI	Basic Rate Interface	GR	Generic Requirement		Channels
C	Conditional	GR-815	Generic Requirements For	PRI	Primary Rate Interface
CAS	Channel Associated Signaling		Network Element/Network System (NE/NS) Security	PSTN	Public Switched Telephone Network
CJCSI	Chairman of the Joint	H.320	Standard for Narrowband VTC	Q.931	Signaling Standard for ISDN
	Chiefs of Staff Instruction	ISDN	Integrated Services Digital	R	Required
DIACAP	DoD Information		Network	S/T	ISDN BRI 4-wire interface
	Assurance Certification and	IT	Information Technology	STE	Secure Terminal Equipment
	Accreditation Process	ITU-T	International Telecommunication	STIGs	Security Technical
DISR	DoD IT Standards Registry		Union - Telecommunication		Implementation Guides
DoD	Department of Defense		Standardization Sector	STU-III	Secure Telephone Unit -3rd
DoDI	DoD Instruction	kbps	kilobits per second		generation
DP	Dial Pulse	Mbps	Megabits per second	T1	Digital Transmission Link Level 1
DS0	Digital Signal Level 0	MFR1	Multi-Frequency		(1.544 Mbps)
DS1	Digital Signal Level 1		Recommendation 1	T1.607	ISDN Layer 3 Signaling
	(1.544 Mbps) (2.048 Mbps	MLPP	Multi-Level Precedence and		Specification for Circuit Switched
	European)		Preemption		Bearer Service for DSS1
DSN	Defense Switched Network	MOS	Mean Opinion Score	T.4	Standardization of Group 3
DSS1	Digital Subscriber	NI 1/2	National ISDN Standard 1 or 2		facsimile terminals for document
	Signaling 1	NX56	Data format restricted to multiples		transmission
DTMF	Dual Tone Multi-Frequency		of 56 kbps	UCR	Unified Capabilities Requirement
E1	European Basic Multiplex	NX64	Data format restricted to multiples	VBD	Variable bit data
	Rate (2.048 Mbps)		of 64 kbps	VoIP	Voice over Internet Protocol
E911	Enhanced 911 Service	PBX	Private Branch Exchange	VTC	Video Teleconferencing
E&M	Ear and Mouth	PBX2	Private Branch Exchange 2		Ç

5. The SUT system configurations for software version 4.0 SU2 are listed in Table 4. The SUT was verification and validation tested in an operationally realistic environment to determine interoperability with the Siemens EWSD DSN switch noted in Table 4. The Siemens DSN switch tested is not intended to identify the only switch certified with the SUT. The SUT is certified with switching systems listed on the Unified Capabilities (UC) Approved Products List (APL) that offer the same certified interfaces. Figure 1 depicts a notional test diagram. Figure 2 depicts the SUT CIC Software Version 4.0 SU2 test diagram.

Table 4. SUT CIC Software Version 4.0 SU2 Tested System Configurations

System Name		Equipment				
Required Ancillary	Active Directory					
Equipment	Public Key Infrastructure					
Equipment		SysLog Server	:			
System Name		Software Rele	ase			
Siemens EWSD		19d with Patch Se	t 46			
		Card Name				
	Hardware	Part Number/Name	Software/Firmware (see note.)			
		1 at t Number/Name	6.20A.059			
	Audiocodes M800 T1 PRI	NA	Embedded Linux Kernel v2.6.21.7			
			6.20A.055.001			
	Audiocodes M3000 PRI	NA	Embedded Linux Kernel v2.6.21.7			
			IC 4.0.2			
			Internet Explorer 9			
	CIC 4.0 Primary	HP Proliant DL 380 G6	Microsoft .Net Framework 4.0			
	CIC 4.0 I Illiary	In Honant BE 300 G0	MSXML 6.20.2003.0 SP2			
			Windows Server 2008 R2 SP1			
			IC 4.0.2			
			Internet Explorer 9			
	CIC 4.0 Secondary	HP Proliant DL 380 G5	Microsoft .Net Framework 4.0			
			MSXML 6.20.2003.0 SP2			
			Windows Server 2008 R2 SP1			
			Microsoft SQL Server 2008 R2			
	Database Server	HP Server DL 320 G5	(64 bit) 10.50.2500.0			
			MDAC 3.86.3959			
			Internet Explorer 9			
			Microsoft .Net Framework 3.5 SP1			
			MSXML 6.20.2003.0 SP2			
			Windows Server 2008 R2 SP1			
			Solaris 10 11/06 s10s_u3wos_10 SPARC			
			Oracle 11.1.0.7.0			
Interactive Intelligence			Java 1.6.0_31			
CIC Rel. 4.0 SU2			OpenSSL (core Solaris) 0.9.7d			
	Audiocodes EMS Server	Sun-Fire-V215	Open SSL (open source) 1.0.0f			
	radiocodes Elvis server	Sun The V213	Reflection for Secure IT 7.2.0.115			
			Solaris Kerberos			
			(based on MIT Kerberos 5 release 1.4.0)			
			Apache 2.2.21			
			EMS 6.2.94			
			Media Server 4.2.17.1056			
	Madia Saman 4.0 (-2)	IID D1:+ DI 260 C5	Internet Explorer 9 Microsoft .Net Framework 4.0			
	Media Server 4.0 (x2)	HP Proliant DL 360 G5	MSXML 6.20.2003.0 SP2			
			Windows Server 2008 R2 SP1			
			Interaction Client .NET Edition v4.0 SU 2			
			IC User Application 4.02.1056			
			IC Business Manager Applications 4.02.1056			
	Windows 7		ActivClient 6.2.0.50			
	Application Workstation	NA	Desktop Validator SE 4.10.0.344			
	(site-provided)		MSXML 4.0 SP2			
			Microsoft .Net Framework 4.0			
			Microsoft Windows 7 SP1			
			Audiocodes EMS 6.2.94			
			ActivClient 6.2.0.50			
	Windows XP Pro		Desktop Validator SE 4.10.0.344			
	Management Workstation	NA	MSXML 4.0 SP2			
	(site-provided)		Microsoft .Net Framework 3.5 SP1			
			Microsoft Windows XP Pro SP3			

 Table 4. SUT CIC Software Version 4.0 SU2 Tested System Configurations (continued)

System Name	Hardware	Card Name	Sof	ftware/Firmware
	Audiocodes M800 FXS	NA		6.20A.059
Interactive Intelligence	Audiocodes Mood FAS	NA	Embedded Linux Kernel v2.6.21.7	
CIC Rel. 4.0 SU2	Audiocodes MP124	NA		6.20A.055.001
	Audiocodes Wii 124	IVA	pSOS 2.5.4	
		Telephones		
Telephone Type		Model		
	Poly	Polycom SoundPoint 300 series		
IP ¹	Poly	Polycom SoundPoint 400 series Polycom SoundPoint 500 series w/ extension module		
11	Polycom Sound			

Polycom SoundPoint 600 series w/ extension module

Generic

NA

NOTE:

Analog

NOTE: The hardware and software applications bolded and underlined depict the changes from the baseline Interoperability Certification.

LEGEND:			
CIC	Contact Interaction Center		
EMS	Element Management System	OpenSSL	Open Secure Sockets Layer
EWSD	Elektronisches Wählsystem Digital	PRI	Primary Rate Interface
DL	Product Name	Pro	Professional
FXS	Foreign Exchange Station	pSOS	Provably Secure Operating System
HMP	Host Media Processing	R	Release
HP	Hewlett-Packard	Rel.	Release
ININ	Interactive Intelligence Corp	SE	Standard Edition
IC	Interaction Center	SP	Service Pack
IP	Internet Protocol	SPARC	Sun Microsystems Architecture
M	Mediant	SQL	Structured Query Language
Mbps	Megabits per second	SU	Service Update
MDAC	Microsoft Data Access Components	T1	Digital Transmission Link Level 1 (1.544 Mbps)
MSXML	Microsoft Extensible Markup Language	v	version
NA	Not Applicable	XP	Experience

^{1.} The SUT is IPv6 dual stack capable; however, the IP end instruments only support IPv4. DISA adjudicated this as minor on 22 May 2012 with the vendor's POA&M to support IPv6 EIs in new software update targeted for release in December 2012. The SUT will be subject to removal from the APL if the POA&M is not met.

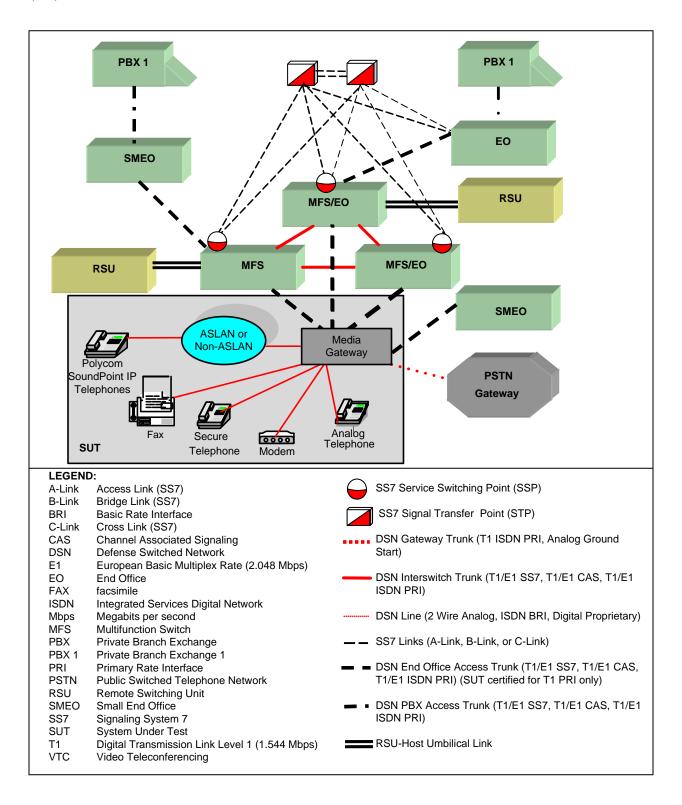


Figure 1. SUT Notional Test Configuration

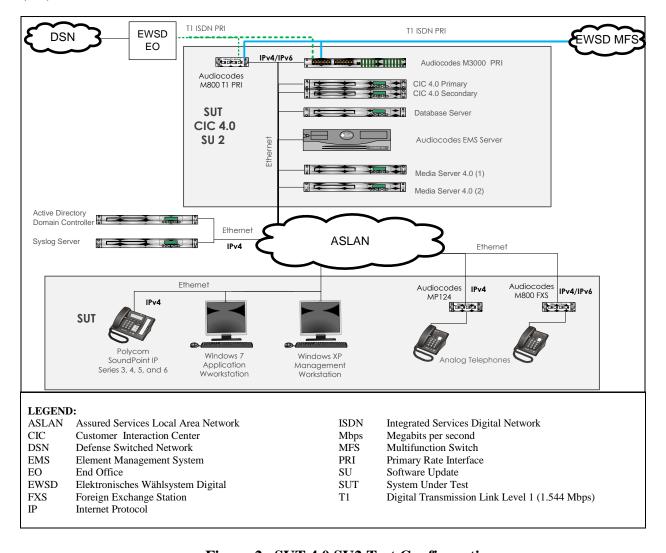


Figure 2. SUT 4.0 SU2 Test Configuration

6. No detailed test report was developed in accordance with the Program Manager's request. JITC distributes interoperability information via the JITC Electronic Report Distribution (ERD) system, which uses Unclassified-But-Sensitive Internet Protocol Router Network (NIPRNet) email. More comprehensive interoperability status information is available via the JITC System Tracking Program (STP). The STP is accessible by .mil/gov users on the NIPRNet at https://stp.fhu.disa.mil. Test reports, lessons learned, and related testing documents and references are on the JITC Joint Interoperability Tool (JIT) at http://jit.fhu.disa.mil (NIPRNet). Information related to DSN testing is on the Telecom Switched Services Interoperability (TSSI) website at http://jitc.fhu.disa.mil/tssi. Due to the sensitivity of the information, the Information Assurance Accreditation Package (IAAP) that contains the approved configuration and deployment guide must be requested directly through government civilian or uniformed military personnel from the Unified Capabilities Certification Office (UCCO), e-mail: disa.meade.ns.list.unified-capabilities-certification-office@mail.mil.

7. The JITC point of contact is Ms. Anita Mananquil, DSN 879-5164, commercial (520) 538-5164, FAX DSN 879-4347, or e-mail to anita.l.mananquil.civ@mail.mil. The JITC's mailing address is P.O. Box 12798, Fort Huachuca, AZ 85670-2798. The tracking number for the SUT is 0918002.

FOR THE COMMANDER:

Enclosure a/s

for RICHARD A. MEADOR

Chief

Battlespace Communications Portfolio

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Medical Health Systems, JMIS IV&V

HQUSAISEC, AMSEL-IE-IS

UCCO

ADDITIONAL REFERENCES

- (c) Chairman of the Joint Chiefs of Staff Instruction (CJCSI) 6215.01C, "Policy for Department of Defense Voice Services with Real Time Services (RTS)," 9 November 2007
- (d) Office of the Assistant Secretary of Defense, "Department of Defense Unified Capabilities Requirements 2008," 22 January 2009
- (e) Joint Interoperability Test Command, Memo, JTE, "Special Interoperability Test Certification of Interactive Intelligence[®] Inc. Customer Interaction CenterTM (CIC) Release 3.0 Software Update (SU) 9," 6 May 2011
- (f) Joint Interoperability Test Command, Memo, "Information Assurance (IA) Assessment of Interactive Intelligence Corporation (ININ) Customer Interaction Center (CIC) Release (Rel.) 3.0 (TN0918002)," 6 April 2011
- (g) Joint Interoperability Test Command, "Defense Switched Network Generic Switch Test Plan (GSTP), Change 2," 2 October 2006